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## NOTIFICATION OF ELECTION

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<b>Applicant</b> THELEN, Eric et al	

1. The designated Office is hereby notified of its election made:

☒ in the demand filed with the International Preliminary Examining Authority on:

23 October 2000 (23.10.00)

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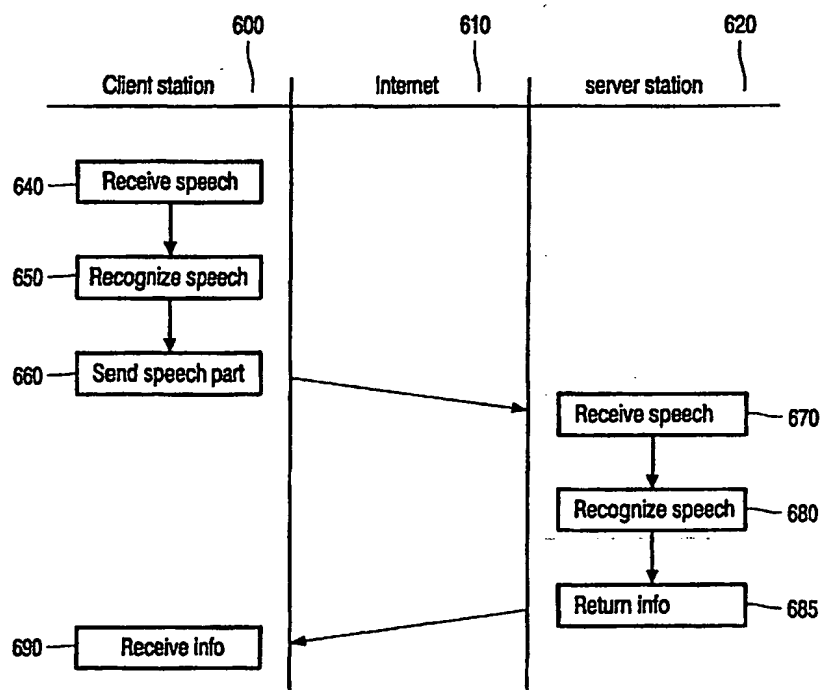
## INTERNATIONAL APPLICATION PUBLISHED UNDER THE PATENT COOPERATION TREATY (PCT)

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(21) International Application Number: <b>PCT/EP00/01975</b> (22) International Filing Date: <b>7 March 2000 (07.03.00)</b> (30) Priority Data: 99200950.6                      26 March 1999 (26.03.99)                      EP 99203342.3                      12 October 1999 (12.10.99)                      EP (71) Applicant (for all designated States except US): <b>KONINKLIJKE PHILIPS ELECTRONICS N.V. [NL/NL]; Groenewoudseweg 1, NL-5621 BA Eindhoven (NL).</b> (71) Applicant (for DE only): <b>PHILIPS CORPORATE INTELLECTUAL PROPERTY GMBH [DE/DE]; Habsburgerallee 11, D-52066 Aachen (DE).</b> (72) Inventors; and (75) Inventors/Applicants (for US only): <b>THELEN, Eric [DE/US]; Prof. Holstlaan 6, NL-5656 AA Eindhoven (NL). BESLING, Stefan [DE/US]; Prof. Holstlaan 6, NL-5656 AA Eindhoven (NL).</b> (74) Agent: <b>GROENENDAAL, Antonius, W., M.; Internationaal Octrooibureau B.V., Prof. Holstlaan 6, NL-5656 AA Eindhoven (NL).</b>		(81) Designated States: <b>AE, AL, AM, AT, AU, AZ, BA, BB, BG, BR, BY, CA, CH, CN, CR, CU, CZ, DE, DK, DM, EE, ES, FI, GB, GD, GE, GH, GM, HR, HU, ID, IL, IN, IS, JP, KE, KG, KP, KR, KZ, LC, LK, LR, LS, LT, LU, LV, MA, MD, MG, MK, MN, MW, MX, NO, NZ, PL, PT, RO, RU, SD, SE, SG, SI, SK, SL, TJ, TM, TR, TT, TZ, UA, UG, US, UZ, VN, YU, ZA, ZW, ARIPO patent (GH, GM, KE, LS, MW, SD, SL, SZ, TZ, UG, ZW), Eurasian patent (AM, AZ, BY, KG, KZ, MD, RU, TJ, TM), European patent (AT, BE, CH, CY, DE, DK, ES, FI, FR, GB, GR, IE, IT, LU, MC, NL, PT, SE), OAPI patent (BF, BJ, CF, CG, CI, CM, GA, GN, GW, ML, MR, NE, SN, TD, TG).</b>  Published With international search report.	

(54) Title: **CLIENT-SERVER SPEECH RECOGNITION**

## (57) Abstract

A distributed speech recognition system (300) includes at least one client station (330) and a server station (310) connected via the public Internet (350). The client station (330) includes means (331) for receiving the speech input signal from a user. A speech controller (335) directs at least part of the speech input signal to a local speech recognizer (334). The, preferably limited, speech recognizer (334) is capable of recognizing at least part of the speech input, for instance a spoken command for starting full recognition. In dependence on the outcome of the recognition, the speech controller (335) selectively directs a part of the speech input signal via the public Internet (350) to the server station (310). The server station (310) includes means (312) for receiving the speech equivalent signal from the public Internet and a large/huge vocabulary speech recognizer (314) for recognizing the received speech equivalent signal.



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Client-server speech recognition.

The invention relates to a distributed speech recognition system for recognizing a speech input signal; the system including at least one client station and a server station; the client station including means for receiving the speech input signal from a user and means for transferring a signal representative of the received speech to the server station via the public Internet; and the server station including means for receiving the speech equivalent signal from the public Internet and a large/huge vocabulary speech recognizer for recognizing the received speech equivalent signal.

The invention also relates to a method of recognizing a speech input signal in a distributed system including at least one client station and a server station.

US 5,819,220 discloses a client-server speech recognition system, wherein the client station is local to the user and the server is located remotely, accessible via the public Internet. This system is used for providing speech input relating to a Web page. The user provides speech input to the client station which displays a Web page using a conventional browser. The speech may, for instance, be used for specifying a query or for filling information fields (e.g. name, and address) of the page. Usually, the client station receives the speech via a microphone and an A/D converter of an audio card. A representation of the speech is sent to a speech server on the public Internet. This server may be located in or be accessible via a Web server that supplied the Web page. The server may also be accessible via the public Internet at a location independent of the Web server. The server recognizes the speech. The recognition output (e.g. a recognized word sequence) may be sent back to the client station or directly to the Web server. In the known system a powerful speech recognizer can be used in the server which is capable of and optimized for recognizing speech in an Internet environment. For certain applications it will be required that this recognizer can support, to a certain extent, the huge vocabularies which can occur in an Internet environment where a user can access virtually any document on any topic. In the known client-server system the client station has no speech recognizer.

Since in the described system all speech input is directed to the server, the load on the server can get very high. This is particularly the case if the system supports many client stations operating at the same time.

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It is an object of the invention to improve the system and method set forth by reducing the load on the server.

To achieve the object according to the invention, the system is characterized in that the client station includes a local speech recognizer and a speech controller; the speech  
10 controller being operative to direct at least part of the speech input signal to the local speech recognizer and, in dependence on the outcome of the recognition, selectively directing a part of the speech input signal via the public Internet to the server station. By incorporating also a recognizer in the client station, load can be removed from the server. The server can be targeted towards the difficult task of providing high quality recognition of huge vocabulary  
15 speech for possibly many simultaneous users and be relieved from simple tasks which the local recognizer can easily fulfill. Although the tasks may be simple, they can remove a high load from the server and the public Internet, simply by making it unnecessary to send all speech input to the server. Moreover, certain recognition tasks can be performed more effectively in the client than in the server, since the client can have easier access to local  
20 information relevant for the recognition.

As defined in the measure of the dependent claim 2, a simple recognizer is used in the client station. In this way the additional costs and processing load on the client station can be kept low.

As defined in the measure of the dependent claim 3, the local recognizer is used  
25 to detect a spoken activation command. This relieves the central recognizer from continuously having to scan the speech input signals coming from the client stations even if the user is not speaking or if the user is speaking but does not want his/her speech to be recognized. It also relieves the public Internet from unnecessary load.

As defined in the measure of the dependent claim 4, the local recognizer is used  
30 for performing recognition of instructions for control of the local client station. The client station is best suited to determine which local operations are possible (e.g. which menu items can be controlled via voice). Moreover, it is avoided that the speech is sent via the public Internet, and the recognition result is sent back, whereas the local station is equally well or even better suited for performing the recognition task.



As defined in the measure of the dependent claim 5, the client station uses its local recognizer to determine to which speech server the speech signal needs to be sent. Such an approach can efficiently be used in situations where there are several speech recognition servers. An example of this is a Web page with contains several advertisement banners of different companies. Some or all of these companies may have their own speech recognition server, for instance to allow a user to phrase spoken queries. The local recognizer/controller may perform the selection of the server and the routing of the speech based on spoken explicit routing commands, such as "select Philips", or "speak to Philips". Information used for recognizing the routing command may be extracted from the banner itself. Such information may be in the banner in the form of a tag, and may include items, such as a textual and phonetic representation of the routing command. The local recognizer/controller may also determine the routing based on information associated with the respective speech server. For instance, words of the banner text may be used as the basis for the routing. For instance, if the user speaks a word which occurs in one of the banners, the speech is directed to the speech server associated with that banner. If a word occurs in more than one banner, the speech may be routed to several speech servers, or to one server which was most likely (e.g. whose associated banner had the highest relative occurrence of the word). Instead of using the words which are explicitly shown in the banner, the banner may also be associated with textual information, e.g. via a link. If the user speaks one or more words from that information, the speech server for the banner is selected.

As defined in the measure of the dependent claim 6, the speech recognizer in the server is used as a kind of 'backup' for those situations in which the local recognizer is not capable of recognizing the user input adequately. The decision to transfer the speech input to the server may be based on performance indications like scores or confidence measures. In this way a conventional large vocabulary recognizer can be used in the client station, whereas a more powerful recognizer is used in the server. The recognizer in the server may, for instance, support a larger vocabulary or more specific language models. The local recognizer may remain operational and recognize the input, even if in parallel the input is also recognized by the server. In this way, the input of the user can still be recognized in 'real time'. The initial recognition of the local recognizer with a possibly lower accuracy can then be replaced by a possibly higher quality result of the server. A selector makes the final choice between the recognition result of the local recognizer and the remote recognizer. This selection may be based on the performance indicators.

To meet the object according to the invention, the method of recognizing a speech input signal in a distributed system includes:

receiving in the client station the speech input signal from a user;  
recognizing at least part of the speech input signal in the client station;  
5 selectively directing a signal representative of a part of the speech input signal via the public Internet from the client station to the server station in dependence on the outcome of the recognition;  
receiving the speech equivalent signal in the server station from the public Internet; and  
10 recognizing the received speech equivalent signal in the server station using a large/huge vocabulary speech recognizer.

These and other aspects of the invention will be apparent from and elucidated with reference to the embodiments shown in the drawings.

Figure 1 shows the elements of a typical speech recognizer;

Figure 2 illustrates HMM-based word models;

Figure 3 shows a block diagram of a distributed speech recognition system according to the invention;

20 Figure 4 illustrates in more detail using the public Internet for access to the server station;

Figure 5 shows a block diagram of a system with a choice of server stations; and

25 Figure 6 illustrates the method of recognizing a speech input signal in a distributed system.

Speech recognition systems, such as large vocabulary continuous speech recognition systems, typically use a collection of recognition models to recognize an input pattern. For instance, an acoustic model and a vocabulary may be used to recognize words and  
30 a language model may be used to improve the basic recognition result. Figure 1 illustrates a typical structure of a large vocabulary continuous speech recognition system 100 [refer L.Rabiner, B-H. Juang, "Fundamentals of speech recognition", Prentice Hall 1993, pages 434 to 454]. The system 100 comprises a spectral analysis subsystem 110 and a unit matching

subsystem 120. In the spectral analysis subsystem 110 the speech input signal (SIS) is spectrally and/or temporally analyzed to calculate a representative vector of features (observation vector, OV). Typically, the speech signal is digitized (e.g. sampled at a rate of 6.67 kHz.) and pre-processed, for instance by applying pre-emphasis. Consecutive samples are grouped (blocked) into frames, corresponding to, for instance, 32 msec. of speech signal. Successive frames partially overlap, for instance, 16 msec. Often the Linear Predictive Coding (LPC) spectral analysis method is used to calculate for each frame a representative vector of features (observation vector). The feature vector may, for instance, have 24, 32 or 63 components. The standard approach to large vocabulary continuous speech recognition is to assume a probabilistic model of speech production, whereby a specified word sequence  $W = w_1 w_2 w_3 \dots w_q$  produces a sequence of acoustic observation vectors  $Y = y_1 y_2 y_3 \dots y_T$ . The recognition error can be statistically minimized by determining the sequence of words  $w_1 w_2 w_3 \dots w_q$  which most probably caused the observed sequence of observation vectors  $y_1 y_2 y_3 \dots y_T$  (over time  $t=1, \dots, T$ ), where the observation vectors are the outcome of the spectral analysis subsystem 110. This results in determining the maximum a posteriori probability:

$$\max P(W|Y), \text{ for all possible word sequences } W$$

By applying Bayes' theorem on conditional probabilities,  $P(W|Y)$  is given by:

$$P(W|Y) = P(Y|W).P(W)/P(Y)$$

Since  $P(Y)$  is independent of  $W$ , the most probable word sequence is given by:

$$\arg \max P(Y | W).P(W) \text{ for all possible word sequences } W \quad (1)$$

In the unit matching subsystem 120, an acoustic model provides the first term of equation (1). The acoustic model is used to estimate the probability  $P(Y|W)$  of a sequence of observation vectors  $Y$  for a given word string  $W$ . For a large vocabulary system, this is usually performed by matching the observation vectors against an inventory of speech recognition units. A speech recognition unit is represented by a sequence of acoustic references. Various forms of speech recognition units may be used. As an example, a whole word or even a group of words may be represented by one speech recognition unit. A word model (WM) provides for each word of a given vocabulary a transcription in a sequence of acoustic references. In most small vocabulary speech recognition systems, a whole word is represented by a speech recognition unit, in which case a direct relationship exists between the word model and the speech recognition unit. In other small vocabulary systems, for instance used for recognizing a relatively large number of words (e.g. several hundreds), or in large vocabulary systems, use can be made of linguistically based sub-word units, such as phones, diphones or syllables, as well as derivative units, such as fenenes and fenones. For such

systems, a word model is given by a lexicon 134, describing the sequence of sub-word units relating to a word of the vocabulary, and the sub-word models 132, describing sequences of acoustic references of the involved speech recognition unit. A word model composer 136 composes the word model based on the subword model 132 and the lexicon 134.

5           Figure 2A illustrates a word model 200 for a system based on whole-word speech recognition units, where the speech recognition unit of the shown word is modeled using a sequence of ten acoustic references (201 to 210). Figure 2B illustrates a word model 220 for a system based on sub-word units, where the shown word is modeled by a sequence of three sub-word models (250, 260 and 270), each with a sequence of four acoustic references  
10           (251, 252, 253, 254; 261 to 264; 271 to 274). The word models shown in Fig. 2 are based on Hidden Markov Models (HMMs), which are widely used to stochastically model speech signals. Using this model, each recognition unit (word model or subword model) is typically characterized by an HMM, whose parameters are estimated from a training set of data. For large vocabulary speech recognition systems usually a limited set of, for instance 40, sub-word  
15           units is used, since it would require a lot of training data to adequately train an HMM for larger units. An HMM state corresponds to an acoustic reference. Various techniques are known for modeling a reference, including discrete or continuous probability densities. Each sequence of acoustic references which relate to one specific utterance is also referred as an acoustic transcription of the utterance. It will be appreciated that if other recognition  
20           techniques than HMMs are used, details of the acoustic transcription will be different.

          A word level matching system 130 of Fig. 1 matches the observation vectors against all sequences of speech recognition units and provides the likelihoods of a match between the vector and a sequence. If sub-word units are used, constraints can be placed on the matching by using the lexicon 134 to limit the possible sequence of sub-word units to  
25           sequences in the lexicon 134. This reduces the outcome to possible sequences of words.

          Furthermore a sentence level matching system 140 may be used which, based on a language model (LM), places further constraints on the matching so that the paths investigated are those corresponding to word sequences which are proper sequences as specified by the language model. As such the language model provides the second term  $P(W)$   
30           of equation (1). Combining the results of the acoustic model with those of the language model, results in an outcome of the unit matching subsystem 120 which is a recognized sentence (RS) 152. The language model used in pattern recognition may include syntactical and/or semantical constraints 142 of the language and the recognition task. A language model based on syntactical constraints is usually referred to as a grammar 144. The grammar 144 used by

the language model provides the probability of a word sequence  $W = w_1 w_2 w_3 \dots w_q$ , which in principle is given by:

$$P(W) = P(w_1)P(w_2|w_1).P(w_3|w_1w_2)\dots P(w_q|w_1w_2w_3\dots w_q).$$

Since in practice it is infeasible to reliably estimate the conditional word probabilities for all words and all sequence lengths in a given language, N-gram word models are widely used. In an N-gram model, the term  $P(w_j|w_1w_2w_3\dots w_{j-1})$  is approximated by  $P(w_j|w_{j-N+1}\dots w_{j-1})$ . In practice, bigrams or trigrams are used. In a trigram, the term  $P(w_j|w_1w_2w_3\dots w_{j-1})$  is approximated by  $P(w_j|w_{j-2}w_{j-1})$ .

Figure 3 shows a block diagram of a distributed speech recognition system 300 according to the invention. Examples of the working of the system will be described in particular for an application wherein recognized speech is converted to a textual or similar representation. Such a textual representation may be used for dictation purposes, where the textual representation is entered in a document, e.g. in word processor, or in a textual field, e.g. for specifying a field in a database. For dictation, current large vocabulary recognizers support an active vocabulary and lexicon of up to 60,000 words. It is difficult to obtain sufficient relevant data to build models capable of sufficiently accurate recognition for a much larger number of words. Typically, a user may add a certain number of words to the active vocabulary/lexicon. Those words can be retrieved from a background vocabulary of 300,000 to 500,000 words (which also contains an acoustic transcription of the words). For dictation or similar purposes a huge vocabulary may, for instance, consist of at least 100,000 active words or even over 300,000 active words. It will be appreciated that particularly for an Internet environment where by a click on a link an entirely different context may be created it is preferred that many of the words of the background vocabulary can be actively recognized. For other recognition tasks, such as recognizing names, which are usually modeled as a flat list with some form of prior name probability attached to it, but for which no high quality language model exists, a vocabulary of over 50,000 words can already be classified as huge.

It will be understood that the recognition outcome need not be used for dictation purposes. It may equally well be used as input for other systems, such as dialogue systems, wherein depending on the recognized speech information is retrieved from a database or an operation is effected, like ordering a book or reserving a journey.

The distributed recognition system 300 comprises a server station 310 and at least one client station. Shown are three client stations 320, 330 and 340, where further details are only shown for client station 330. The stations may be implemented using conventional computer technology. For instance, the client station 330 may be formed by a desk-top

personal computer or workstation, whereas the server station 310 may be formed by a PC server or workstation server. The computers are operated under control of a suitable program loaded in the processor of the computer. The server station 310 and the client stations 320, 330, and 340 are connected via the public Internet 350.

5 Figure 4 illustrates in more detail the use of the public Internet 400. The server station 310 can be seen as a service provider which provides a service to users (client stations) via the public Internet. The service offered by the service provider may be restricted to speech-to-text conversion (receiving the speech via the public Internet and returning the recognition outcome in a textual or similar form). The service provider may also offer enhanced  
10 functionality, such as acting as a portal to various types of information available in the public Internet. Such enhanced functionality may rely on the speech recognized by the server station, in which case the recognized speech need not be returned to the client station. The client stations get access to the public Internet 400 via access providers. Shown are two access providers 410 and 420. In the example, access provider 410 provides access to the client  
15 station 320, whereas the access provider 420 provides access to the client stations 330 and 340. The connection between the access provider and its client station(s) usually occurs via a wide area network, like a dial-in telephone connection or a cable TV connection.

As illustrated in Fig.3, the stations comprise communication means 312 and 332, respectively, for communicating via the public Internet 350. Any communication means  
20 suitable for use in combination with the public Internet 350 may be used. Typically, the communication means are formed by a combination of hardware, such as a communication interface or a modem, and software in the form of a software driver supporting a specific communication protocol, including Internet's TCP/IP protocols. The client station 330 comprises means for receiving speech from a user, for instance via an interface 331. The client  
25 station 330 further comprises means for pre-processing the speech signal, making it suitable for transfer to the server station 310. For instance, the client station may comprise a spectral analysis subsystem 333 similar to the spectral analysis subsystem 110 of Fig.1. The server station 310 is capable of performing all other tasks as described for system 100 of Fig.1. Preferably, the server station is capable of large or huge vocabulary continuous speech  
30 recognition. In itself large vocabulary speech recognition is well-known. For huge vocabulary speech recognition preferably the technique described in the European application EP 99200949.8 is used. The server station 310 comprises a recognizer 312 which may be similar to the unit matching subsystem of Fig.1, using a word model and a language model. The recognition output, such as a sequence of recognized words RS, may be transmitted back to

the client station 330. It may also be used in the server station 310 for further processing (e.g. in a dialogue system), or be transferred to a further station, like a service provider on Internet. According to the invention, the client station 330 includes a speech recognizer 334. In a preferred embodiment, the speech recognizer 334 of the client station 330 is a limited speech recognizer, such as a small vocabulary or keyword recognizer. Such recognizers are well-known and can be implemented cost-effectively. The client station 330 further includes a speech controller 335. The speech controller 335 analyzes the outcome of the recognition by the local recognizer 334 and, in dependence on the outcome of the recognition, selectively directs a part of the speech input signal via the public Internet 350 to the server station 310. To this end, the client station 330 includes a controllable switch 336 which determines whether or not the speech input received via the interface 331 (and analyzed by the analyzer 333) is sent by means of the communication interface 332 to the server station 310. Depending on the task of the local recognizer 334 and the load which may be put on the client station 330, the speech controller 335 may use a further switch to only transfer part of the speech input signal to the local speech recognizer 334. For instance, the client station 330 may include a speech activity detector which, for instance based on the energy level of the received input signal, determines whether or not the user might be speaking. If the user is not speaking (the level is below a threshold), the signal need not be directed to the local recognizer 334. Such activity detectors are known. If the load is not a limiting factor, it is preferred to always direct the speech to the local recognizer, at least when the user has activated the involved programs and as such has indicated a wish to sometime start a recognition session. By always having the local recognizer active, defects of the energy-based activity detectors can be avoided. For instance, it is very difficult to achieve an accurate setting of a threshold for such a detector, particularly in a noisy environment, e.g. with background music or people speaking in the background.

In a preferred embodiment, the speech recognizer 334 is a limited speech recognizer operative to recognize a spoken command for activating recognition by the server station. If such a predetermined command (or one of a number of predetermined activation commands) has been recognized, the speech controller 335 controls the switch 336 to direct the speech input signal to the server station 310. The speech controller 335 also sends an activation instruction to the server station 310 via the communication means 332. This activation instruction may be implicit. For instance, recognition in the server 310 may automatically be activated whenever speech is received via the communication interface 312.

In an alternative embodiment, the recognizer 334 is operative to recognize at least one spoken command/control instruction relating to the operation of the local client.

station 330. For this task it is sufficient to use a small vocabulary or keyword recognizer. In response to recognizing a local command/control instruction, the speech controller 335 issues a corresponding machine control instruction to a station controller 337 of the local client station 330. Such a machine instruction may be an operating system or user interface instruction as, for instance, defined for the Windows operating system. The speech controller 335 as well as the station controller 337 may be tasks under Windows.

In an alternative embodiment, as shown in Figure 5, the system 500 includes a plurality of server stations, shown are 510, 520 and 530. The server stations may be of similar construction as described for server station 310 of Fig.3. The server stations are connected via the public Internet 540 to at least one client station, shown is client station 550. The client station 550 may be of a same construction as described for station 330 of Fig.3. The speech recognizer of the client station 550 is, preferably, a limited speech recognizer. The speech recognizer is operative to recognize speech routing commands. The speech controller of the client station selectively directs the speech input signal to at least one server associated with the recognized routing command. The directing may be based on a table for translating a speech routing command to a network address of a server station. Such a table may be predetermined, e.g. once loaded into the client station or entered by the user of the client station. Alternatively, such a table may be dynamic. For instance, the table may be embedded in or associated with a document, such an HTML document, downloaded via Internet. In such a document, different areas may be associated with different respective servers. For instance, a document may included several advertisement banners, each associated with its own recognition server.

In an alternative embodiment, the speech recognizer 334 of the client station 330 of Fig.3 is a large vocabulary speech recognizer (or optionally a huge vocabulary speech recognizer). The speech controller 335 directs part (or all) of the speech input signal to the server station 310 if a performance indicator for a recognition result of the speech recognizer 334 in the local client station 330 is below a predetermined threshold. The directing may be effected by changing the switch 336. In principle it may be sufficient to only route the last part of the speech signal to the server station 310, e.g. only the part relating to a badly recognized sentence. It may be preferred to route also earlier speech material to the server station 310, allowing the server station to better synchronize with the speech signal and, optionally, choose suitable recognition models, such as acoustic or language models based on the earlier part of the signal. The server station 310 transfers a recognized word sequence back to the client station 330. The client station 330 including a selector for selecting a recognized word



sequence from the word sequences respectively recognized by the recognizer in the client station and recognizer in the server station. In practice the task of the selector will be combined with the task of the speech controller 335, which anyhow already analyzes the results of the local recognition.

- 5                Figure 6 illustrates the method of recognizing a speech input signal in a distributed system including at least one client station 600 and a server station 620 able to communicate via the public Internet 610. The method includes step 640 of receiving in the client station 600 the speech input signal from a user. In step 650 at least part of the speech input signal is recognized in the client station 600. In step 660 a signal representative of a part
- 10 of the speech input signal is selectively directed via the public Internet 610 from the client station 600 to the server station 620. The directing depends on the outcome of the recognition in the client station 600. In step 670, the speech equivalent signal is received in the server station 620 from the public Internet 610. In step 680 the received speech equivalent signal is recognized in the server station 620 using a large/huge vocabulary speech recognizer.
- 15                Optionally in step 685, the server station returns information representative of a recognition outcome (e.g. a textual transcription of the recognized speech) of the recognition of the speech representative signal back to the client station 600 via the public Internet 610. In step 690, the client station receives the information.

## CLAIMS:

1. A distributed speech recognition system for recognizing a speech input signal; the system including at least one client station and a server station;  
the client station including means for receiving the speech input signal from a user and means for transferring a signal representative of the received speech to the server station via the public Internet; and  
5 the server station including means for receiving the speech equivalent signal from the public Internet and a large/huge vocabulary speech recognizer for recognizing the received speech equivalent signal;  
characterized in that the client station includes a local speech recognizer and a speech  
10 controller; the speech controller being operative to direct at least part of the speech input signal to the local speech recognizer and, in dependence on the outcome of the recognition, selectively directing a part of the speech input signal via the public Internet to the server station.
- 15 2. The system as claimed in claim 1, wherein the local speech recognizer of the client station is a limited speech recognizer, such as a small vocabulary or keyword recognizer.
- 20 3. The system as claimed in claim 1, wherein the local speech recognizer is operative to recognize a spoken command for activating recognition by the server station; and wherein the speech controller is operative to send, in response to recognizing an activation command, an activation instruction to the server and to direct the speech input signal to the server station.
- 25 4. The system as claimed in claim 1, wherein the recognizer is operative to recognize at least one spoken command/control instruction relating to the operation of the local client station and the speech controller is operative to issue a machine control instruction corresponding to the recognized command/control instruction to a station controller of the local client station.

5. The system as claimed in claim 1, characterized in that the system includes a plurality of server stations; the speech recognizer being operative to recognize speech routing commands; the speech controller being operative to selectively direct the speech input signal to at least one associated server.

5

6. The system as claimed in claim 1, wherein the speech recognizer of the client station is a large vocabulary speech recognizer; the speech controller being operative to direct at least a part of the speech input signal to the server station if a performance indicator for a recognition result of the speech recognizer in the local client station is below a predetermined threshold;

10

the server station being operative to transfer a recognized word sequence back to the client station;

the client station including a selector for selecting a recognized word sequence from the word sequences respectively recognized by the recognizer in the client station and recognizer in the server station.

15

7. A method of recognizing a speech input signal in a distributed system including at least one client station and a server station; the method including:

receiving in the client station the speech input signal from a user;

20

recognizing at least part of the speech input signal in the client station;

selectively directing a signal representative of a part of the speech input signal via the public Internet from the client station to the server station; the directing being in dependence on the outcome of the recognition in the client station;

receiving the speech equivalent signal in the server station from the public

25

Internet; and

recognizing the received speech equivalent signal in the server station using a large/huge vocabulary speech recognizer.

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1/3

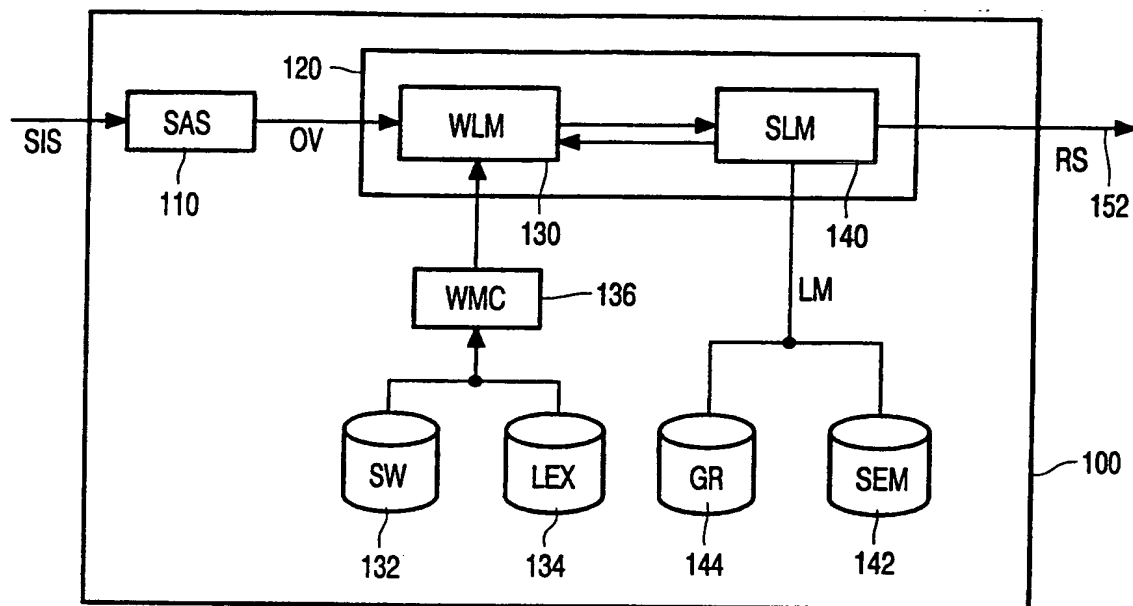


FIG. 1

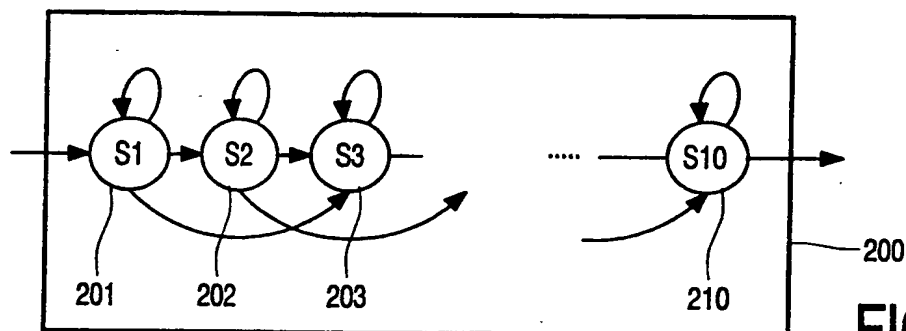


FIG. 2a

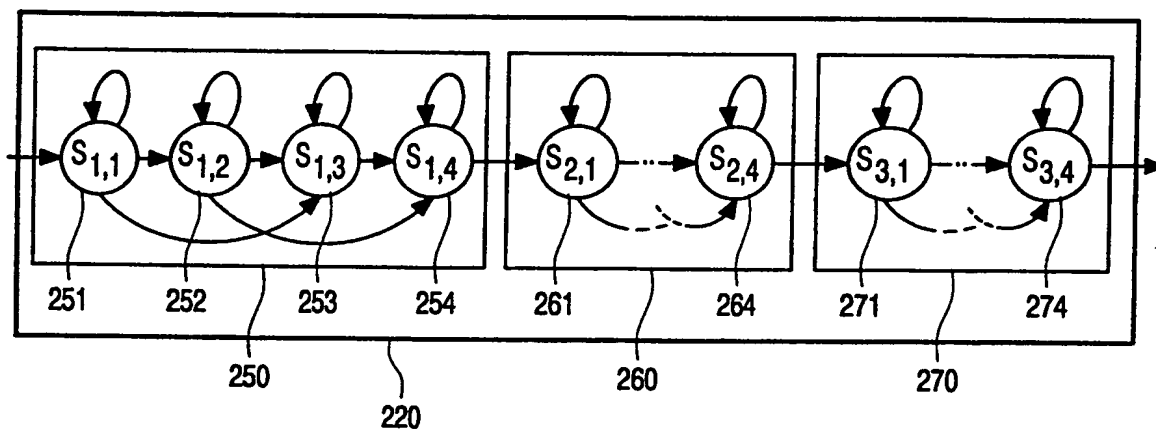


FIG. 2b

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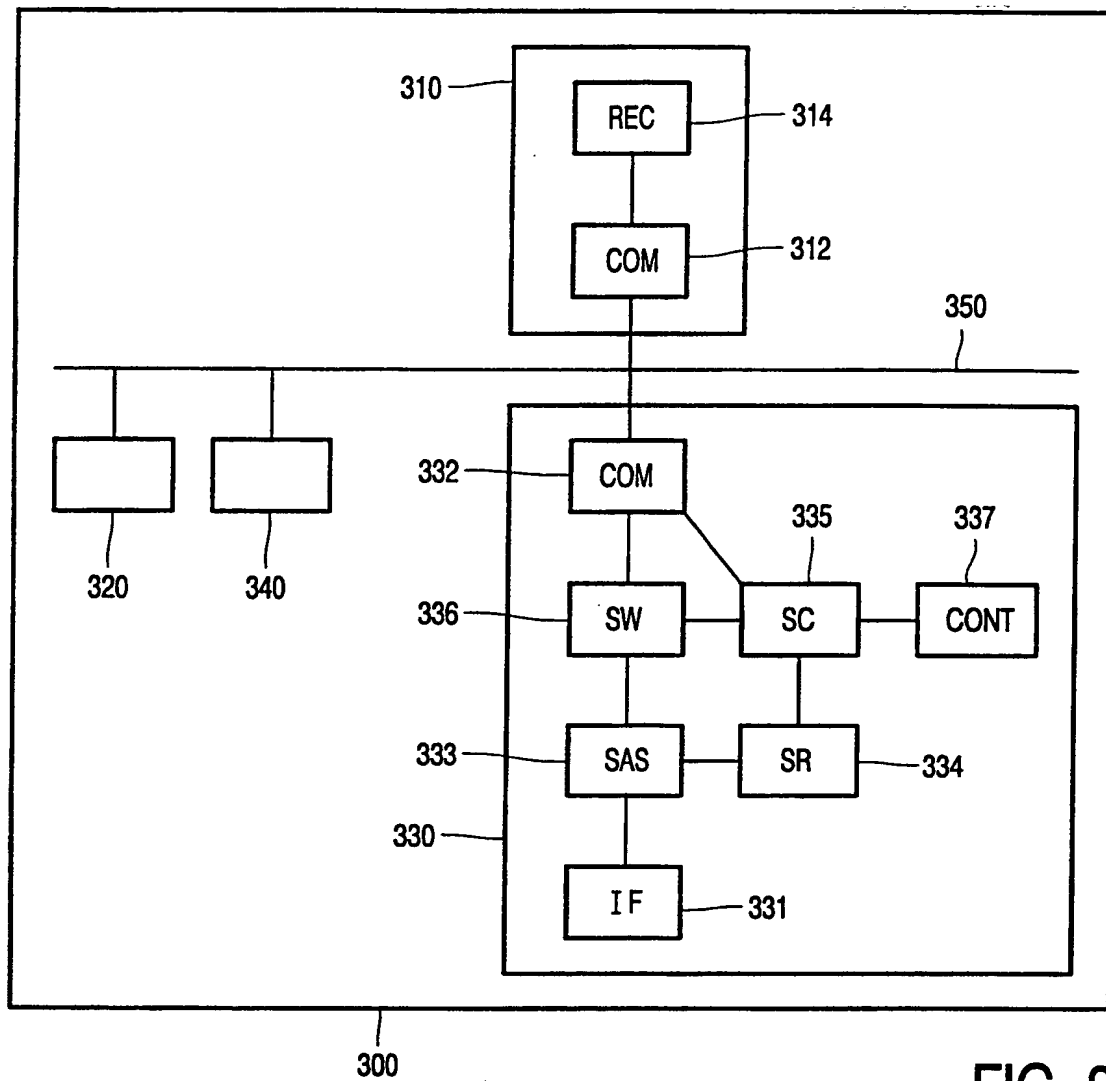


FIG. 3

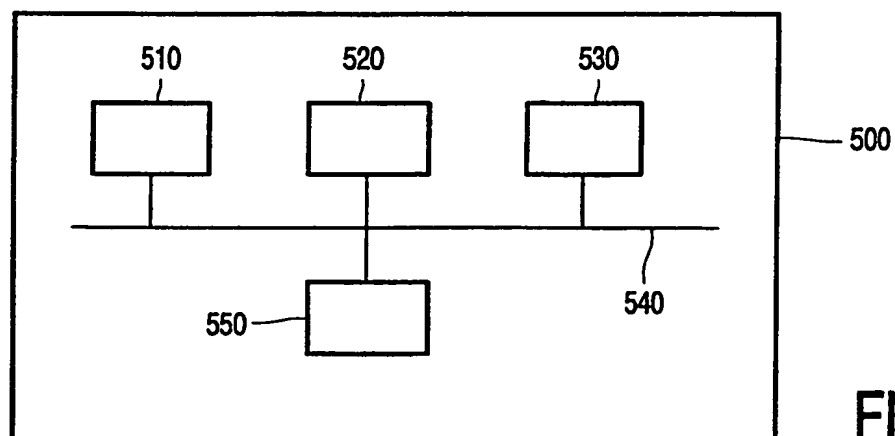


FIG. 5

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3/3

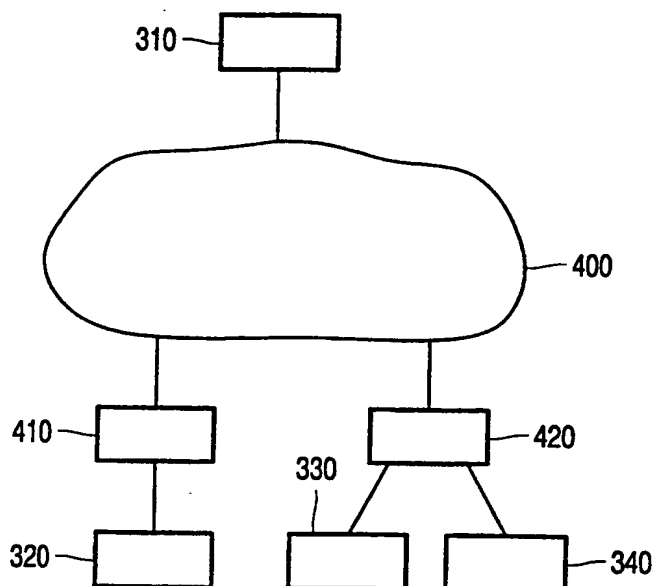


FIG. 4

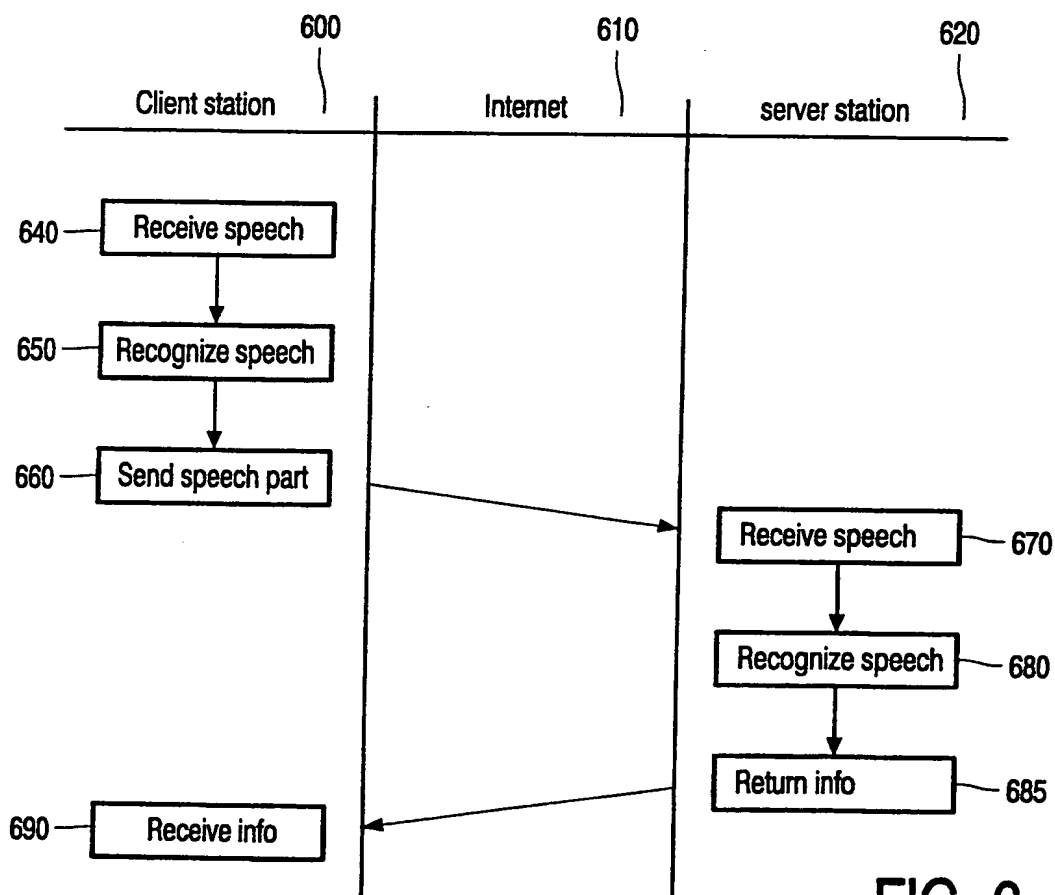


FIG. 6

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# INTERNATIONAL SEARCH REPORT

International Application No

PCT/EP 00/01975

## A. CLASSIFICATION OF SUBJECT MATTER

IPC 7 G10L15/26

According to International Patent Classification (IPC) or to both national classification and IPC

## B. FIELDS SEARCHED

Minimum documentation searched (classification system followed by classification symbols)

IPC 7 G10L

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched

Electronic data base consulted during the international search (name of data base and, where practical, search terms used)

## C. DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
X	WO 98 34217 A (DRAGON SYSTEMS INC) 6 August 1998 (1998-08-06)	1,2,6
Y	abstract; figures 3,4,7 page 3, line 21 -page 4, line 11 page 5, line 11-24 page 11, line 5-33 page 12, line 11-26	3-5
X	WO 95 17746 A (QUALCOMM INC) 29 June 1995 (1995-06-29) abstract; figures 2,5 page 5, line 8-19 page 6, line 6-25 page 8, line 22-27 page 10, line 29 -page 11, last line	1,2
	-/--	



Further documents are listed in the continuation of box C.



Patent family members are listed in annex.

### \* Special categories of cited documents :

"A" document defining the general state of the art which is not considered to be of particular relevance

"E" earlier document but published on or after the international filing date

"L" document which may throw doubts on priority claim(s) or which is cited to establish the publication date of another citation or other special reason (as specified)

"O" document referring to an oral disclosure, use, exhibition or other means

"P" document published prior to the international filing date but later than the priority date claimed

"T" later document published after the international filing date or priority date and not in conflict with the application but cited to understand the principle or theory underlying the invention

"X" document of particular relevance; the claimed invention cannot be considered novel or cannot be considered to involve an inventive step when the document is taken alone

"Y" document of particular relevance; the claimed invention cannot be considered to involve an inventive step when the document is combined with one or more other such documents, such combination being obvious to a person skilled in the art.

"&" document member of the same patent family

Date of the actual completion of the international search

19 June 2000

Date of mailing of the international search report

27/06/2000

Name and mailing address of the ISA

European Patent Office, P.B. 5818 Patentlaan 2  
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Authorized officer

Quélavoine, R

# INTERNATIONAL SEARCH REPORT

Inter- vial Application No  
PCT/EP 00/01975

## C.(Continuation) DOCUMENTS CONSIDERED TO BE RELEVANT

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
Y	<p>"CLIENT-SERVER MODEL FOR SPEECH RECOGNITION" IBM TECHNICAL DISCLOSURE BULLETIN, US, IBM CORP. NEW YORK, vol. 36, no. 3, 1 March 1993 (1993-03-01), pages 25-26, XP000354688 ISSN: 0018-8689 page 25, paragraphs 5,6 page 26, paragraph 4</p>	3-5

# INTERNATIONAL SEARCH REPORT

...information on patent family members

International Application No

PCT/EP 00/01975

Patent document cited in search report	Publication date	Patent family member(s)	Publication date
WO 9834217 A	06-08-1998	EP 0954848 A	10-11-1999
WO 9517746 A	29-06-1995	AU 692820 B	18-06-1998
		AU 1375395 A	10-07-1995
		BR 9408413 A	05-08-1997
		CA 2179759 A	29-06-1995
		CN 1138386 A	18-12-1996
		EP 0736211 A	09-10-1996
		FI 962572 A	20-08-1996
		JP 9507105 T	15-07-1997
		US 5956683 A	21-09-1999
		ZA 9408426 A	30-06-1995

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PCT

INTERNATIONAL PRELIMINARY EXAMINATION REPORT PCT

(PCT Article 36 and Rule 70)

REC'D 06 JUN 2001

WIPCO

Applicant's or agent's file reference PHD 99.144WO	<b>FOR FURTHER ACTION</b> See Notification of Transmittal of International Preliminary Examination Report (Form PCT/IPEA/416)	
International application No. PCT/EP00/01975	International filing date (day/month/year) 07/03/2000	Priority date (day/month/year) 26/03/1999
International Patent Classification (IPC) or national classification and IPC G10L15/26		
Applicant KONINKLIJKE PHILIPS ELECTRONICS N.V.		

1. This international preliminary examination report has been prepared by this International Preliminary Examining Authority and is transmitted to the applicant according to Article 36.


2. This REPORT consists of a total of 6 sheets, including this cover sheet.

- ☒ This report is also accompanied by ANNEXES, i.e. sheets of the description, claims and/or drawings which have been amended and are the basis for this report and/or sheets containing rectifications made before this Authority (see Rule 70.16 and Section 607 of the Administrative Instructions under the PCT).

These annexes consist of a total of 2 sheets.

3. This report contains indications relating to the following items:

- I ☒ Basis of the report
- II ☐ Priority
- III ☐ Non-establishment of opinion with regard to novelty, inventive step and industrial applicability
- IV ☐ Lack of unity of invention
- V ☒ Reasoned statement under Article 35(2) with regard to novelty, inventive step or industrial applicability; citations and explanations supporting such statement
- VI ☐ Certain documents cited
- VII ☒ Certain defects in the international application
- VIII ☐ Certain observations on the international application

Date of submission of the demand  23/10/2000	Date of completion of this report  01.06.2001
Name and mailing address of the international preliminary examining authority:   European Patent Office D-80298 Munich Tel. +49 89 2399 - 0 Tx: 523656 epmu d Fax: +49 89 2399 - 4465	Authorized officer  Greiser, N  Telephone No. +49 89 2399 7402



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**INTERNATIONAL PRELIMINARY  
EXAMINATION REPORT**

International application No. PCT/EP00/01975

**I. Basis of the report**

1. With regard to the **elements** of the international application (*Replacement sheets which have been furnished to the receiving Office in response to an invitation under Article 14 are referred to in this report as "originally filed" and are not annexed to this report since they do not contain amendments (Rules 70.16 and 70.17)*):

**Description, pages:**

1-11 as originally filed

**Claims, No.:**

1-7 as received on 07/05/2001 with letter of 07/05/2001

**Drawings, sheets:**

1/3-3/3 as originally filed

2. With regard to the **language**, all the elements marked above were available or furnished to this Authority in the language in which the international application was filed, unless otherwise indicated under this item.

These elements were available or furnished to this Authority in the following language: , which is:

- ☐ the language of a translation furnished for the purposes of the international search (under Rule 23.1(b)).  
☐ the language of publication of the international application (under Rule 48.3(b)).  
☐ the language of a translation furnished for the purposes of international preliminary examination (under Rule 55.2 and/or 55.3).

3. With regard to any **nucleotide and/or amino acid sequence** disclosed in the international application, the international preliminary examination was carried out on the basis of the sequence listing:

- ☐ contained in the international application in written form.  
☐ filed together with the international application in computer readable form.  
☐ furnished subsequently to this Authority in written form.  
☐ furnished subsequently to this Authority in computer readable form.  
☐ The statement that the subsequently furnished written sequence listing does not go beyond the disclosure in the international application as filed has been furnished.  
☐ The statement that the information recorded in computer readable form is identical to the written sequence listing has been furnished.

4. The amendments have resulted in the cancellation of:

- ☐ the description, pages:  
☐ the claims, Nos.:

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# INTERNATIONAL PRELIMINARY EXAMINATION REPORT

International application No. PCT/EP00/01975

☐ the drawings, sheets:

5. ☐ This report has been established as if (some of) the amendments had not been made, since they have been considered to go beyond the disclosure as filed (Rule 70.2(c)):

*(Any replacement sheet containing such amendments must be referred to under item 1 and annexed to this report.)*

6. Additional observations, if necessary:

## V. Reasoned statement under Article 35(2) with regard to novelty, inventive step or industrial applicability; citations and explanations supporting such statement

### 1. Statement

Novelty (N)	Yes: Claims 1-7
	No: Claims
Inventive step (IS)	Yes: Claims 3,5
	No: Claims 1-2,4,6-7
Industrial applicability (IA)	Yes: Claims 1-7
	No: Claims

2. Citations and explanations  
**see separate sheet**

## VII. Certain defects in the international application

The following defects in the form or contents of the international application have been noted:  
**see separate sheet**

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To Section V:

1. Claim 1 does not meet the requirement of Article 33(3) PCT for the following reason:

Document D1= WO-A-98 34217 (= closest prior art) discloses a distributed speech recognition system for recognizing a speech input signal; a client-server architecture is provided (see p. 3, l. 21-28; and p. 12, l. 11-24) wherein

- a client station includes means for receiving the input signal from a user and means for transferring a signal representative of the received speech to a server station (see p. 10, l. 23-30; and p. 11, l. 5-10);
- the server station includes means for receiving the speech equivalent signal from the client station and a large vocabulary speech recognizer for recognizing the received speech equivalent signal (see p. 11, l. 22-28); and
- the client station includes a local speech recognizer and a speech controller; the speech controller being operative to direct at least part of the speech input signal to the local speech recognizer and, in dependence on a speech routing voice command recognized by the local recognizer, selectively directing a part of the speech input signal to the server station (see p. 5, l. 18-24; p. 18, l. 28- p. 19, l. 6; and p. 11, l. 13-21) [since D1 discloses a software-based selection process, a corresponding speech controller is implicitly assumed. Moreover, the local speech recognizer in D1 comprises "other implementations" which send "voice commands" to the server station. Because of this, D1 implicitly discloses a speech controller which, in dependence of a "speech routing voice command" recognized by the local recognizer, selectively directs a part of the speech input signal to the server station. In other words, said "voice commands" in D1 are a part of the speech input signal].

In D1, the distributed speech recognition system is based on a client-server architecture. It would readily occur to the skilled person that the disclosed distributed speech recognition system could be linked across a network such as the public Internet. Accordingly, it would be obvious to the skilled person that the data transfer between the client station and the remotely-available server station is performed via the public Internet. As a consequence, the subject-matter of claim 1 is obvious to a skilled person and, hence, does not involve an inventive step.

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**INTERNATIONAL PRELIMINARY  
EXAMINATION REPORT - SEPARATE SHEET**

---

International application No. PCT/EP00/01975

2. Claim 2 does not involve an inventive step since D1 discloses a client station wherein a local speech recognizer is a small vocabulary recognizer.
3. Claim 3 is new and appears to involve an inventive step.
4. Claim 4 does not involve an inventive step since D1 discloses a system wherein a local recognizer is operative to recognize a spoken command/control instruction relating to the operation of the local client station (see p. 11, l. 10-18). Since D1 discloses a software-related process, a speech controller is implicitly assumed which issues a machine control instruction corresponding to the recognized command/control instruction to a station controller of the local client station.
5. Claim 5 is new and appears to involve an inventive step.
6. Claim 6 does not involve an inventive step. D1 discloses a distributed speech recognition system wherein the server station comprises a large vocabulary speech recognizer and a selector for selecting a recognized word sequence from the word sequences respectively recognized by the recognizer in the client station and recognizer in the server station (see p. 12, l. 18-22). It would readily occur to the skilled person that the selector in D1 could also be located in the client station. Furthermore, it would be within the competence of the skilled person to adapt the concept of a distributed recognition system of D1 to a system wherein the client station and the server station comprise a large vocabulary speech recognizer and a huge vocabulary speech recognizer, respectively. As a result, the subject-matter of claim 6 is obvious and non-inventive.
7. Claim 7 claims a method of recognizing a speech input in a distributed system. Since method claim 7 corresponds to apparatus claim 1, the same objection of lack of inventive step is raised against claim 7. As a result, the subject-matter of claim 7 does not involve an inventive step.

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**INTERNATIONAL PRELIMINARY  
EXAMINATION REPORT - SEPARATE SHEET**

---

International application No. PCT/EP00/01975

To Section VII:

8. Document D1 indicates background art of the present application. However, neither D1 is cited in the description nor is the relevant prior art disclosed therein briefly discussed (Rule 5.1(ii) PCT).

Technical features mentioned in the claims relating to features depicted in the drawings, are not followed by reference signs placed between parentheses (Rule 6.2(b) PCT).

Remarks:

9. Apparently, the sought scope of protection of the subject-matter of claim 1 is too broad. An amended claim 1 based upon originally filed claim 1 claiming a speech controller, "in dependence on an activation command recognized by the local recognizer, selectively directing a part of the speech input via the public Internet to the server station" (see lines 11-13) is narrower in scope than originally filed claim 1 and, hence, would be new and inventive. In other words, the deletion of the term "or speech routing voice" from the wording of the claim would make the claim new and inventive.

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## CLAIMS:

1. A distributed speech recognition system for recognizing a speech input signal;  
the system including at least one client station and a server station;  
the client station including means for receiving the speech input signal from a  
user and means for transferring a signal representative of the received speech to the server  
5 station via the public Internet; and  
the server station including means for receiving the speech equivalent signal  
from the public Internet and a large/huge vocabulary speech recognizer for recognizing the  
received speech equivalent signal;  
characterized in that the client station includes a local speech recognizer and a speech  
10 controller; the speech controller being operative to direct at least part of the speech input  
signal to the local speech recognizer and, in dependence on the outcome of the recognition,  
selectively directing a part of the speech input signal via the public Internet to the server  
station.
- 15 2. The system as claimed in claim 1, wherein the local speech recognizer of the  
client station is a limited speech recognizer, such as a small vocabulary or keyword  
recognizer.
3. The system as claimed in claim 1, wherein the local speech recognizer is  
20 operative to recognize a spoken command for activating recognition by the server station; and  
wherein the speech controller is operative to send, in response to recognizing an activation  
command, an activation instruction to the server and to direct the speech input signal to the  
server station.
- 25 4. The system as claimed in claim 1, wherein the recognizer is operative to  
recognize at least one spoken command/control instruction relating to the operation of the  
local client station and the speech controller is operative to issue a machine control instruction  
corresponding to the recognized command/control instruction to a station controller of the  
local client station.

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5. The system as claimed in claim 1, characterized in that the system includes a plurality of server stations; the speech recognizer being operative to recognize speech routing commands; the speech controller being operative to selectively direct the speech input signal to at least one associated server.

5 6. The system as claimed in claim 1, wherein the speech recognizer of the client station is a large vocabulary speech recognizer; the speech controller being operative to direct at least a part of the speech input signal to the server station if a performance indicator for a recognition result of the speech recognizer in the local client station is below a predetermined  
10 threshold;

the server station being operative to transfer a recognized word sequence back to the client station;

the client station including a selector for selecting a recognized word sequence from the word sequences respectively recognized by the recognizer in the client station and  
15 recognizer in the server station.

7. A method of recognizing a speech input signal in a distributed system including at least one client station and a server station; the method including:

receiving in the client station the speech input signal from a user;

20 recognizing at least part of the speech input signal in the client station;

selectively directing a signal representative of a part of the speech input signal via the public Internet from the client station to the server station; the directing being in dependence on the outcome of the recognition in the client station;

receiving the speech equivalent signal in the server station from the public  
25 Internet; and

recognizing the received speech equivalent signal in the server station using a large/huge vocabulary speech recognizer.

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## TENT COOPERATION TREATY

## PCT

## INTERNATIONAL SEARCH REPORT

(PCT Article 18 and Rules 43 and 44)

Applicant's or agent's file reference <b>PHD 99.144W0</b>	<b>FOR FURTHER ACTION</b> see Notification of Transmittal of International Search Report (Form PCT/ISA/220) as well as, where applicable, item 5 below.	
International application No. <b>PCT/EP 00/ 01975</b>	International filing date (day/month/year) <b>07/03/2000</b>	(Earliest) Priority Date (day/month/year) <b>26/03/1999</b>
Applicant  <b>KONINKLIJKE PHILIPS ELECTRONICS N.V.</b>		

This International Search Report has been prepared by this International Searching Authority and is transmitted to the applicant according to Article 18. A copy is being transmitted to the International Bureau.

This International Search Report consists of a total of 3 sheets.

☒ It is also accompanied by a copy of each prior art document cited in this report.

**1. Basis of the report**

- a. With regard to the **language**, the international search was carried out on the basis of the international application in the language in which it was filed, unless otherwise indicated under this item.

☐ the international search was carried out on the basis of a translation of the international application furnished to this Authority (Rule 23.1(b)).

- b. With regard to any **nucleotide and/or amino acid sequence** disclosed in the international application, the international search was carried out on the basis of the sequence listing :

☐ contained in the international application in written form.

☐ filed together with the international application in computer readable form.

☐ furnished subsequently to this Authority in written form.

☐ furnished subsequently to this Authority in computer readable form.

☐ the statement that the subsequently furnished written sequence listing does not go beyond the disclosure in the international application as filed has been furnished.

☐ the statement that the information recorded in computer readable form is identical to the written sequence listing has been furnished

2. ☐ **Certain claims were found unsearchable** (See Box I).

3. ☐ **Unity of Invention is lacking** (see Box II).

4. With regard to the **title**,

☒ the text is approved as submitted by the applicant.

☐ the text has been established by this Authority to read as follows:

5. With regard to the **abstract**,

☒ the text is approved as submitted by the applicant.

☐ the text has been established, according to Rule 38.2(b), by this Authority as it appears in Box III. The applicant may, within one month from the date of mailing of this international search report, submit comments to this Authority.

6. The figure of the **drawings** to be published with the abstract is Figure No.

☐ as suggested by the applicant.

☐ because the applicant failed to suggest a figure.

☒ because this figure better characterizes the invention.

6

☐ None of the figures.

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## INTERNATIONAL SEARCH REPORT

International Application No

EP 00/01975

**A. CLASSIFICATION OF SUBJECT MATTER**  
IPC 7 G10L15/26

According to International Patent Classification (IPC) or to both national classification and IPC

**B. FIELDS SEARCHED**

Minimum documentation searched (classification system followed by classification symbols)

IPC 7 G10L

Documentation searched other than minimum documentation to the extent that such documents are included in the fields searched

Electronic data base consulted during the international search (name of data base and, where practical, search terms used)

**C. DOCUMENTS CONSIDERED TO BE RELEVANT**

Category *	Citation of document, with indication, where appropriate, of the relevant passages	Relevant to claim No.
X	WO 98 34217 A (DRAGON SYSTEMS INC) 6 August 1998 (1998-08-06)	1,2,6
Y	abstract; figures 3,4,7 page 3, line 21 -page 4, line 11 page 5, line 11-24 page 11, line 5-33 page 12, line 11-26	3-5
X	WO 95 17746 A (QUALCOMM INC) 29 June 1995 (1995-06-29) abstract; figures 2,5 page 5, line 8-19 page 6, line 6-25 page 8, line 22-27 page 10, line 29 -page 11, last line	1,2

☒ Further documents are listed in the continuation of box C.☒ Patent family members are listed in annex.

## \* Special categories of cited documents :

- "A" document defining the general state of the art which is not considered to be of particular relevance
- "E" earlier document but published on or after the international filing date
- "L" document which may throw doubts on priority claim(s) or which is cited to establish the publication date of another citation or other special reason (as specified)
- "O" document referring to an oral disclosure, use, exhibition or other means
- "P" document published prior to the international filing date but later than the priority date claimed

"T" later document published after the international filing date or priority date and not in conflict with the application but cited to understand the principle or theory underlying the invention

"X" document of particular relevance; the claimed invention cannot be considered novel or cannot be considered to involve an inventive step when the document is taken alone

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Date of the actual completion of the international search

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## INTERNATIONAL SEARCH REPORT

International Application No

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